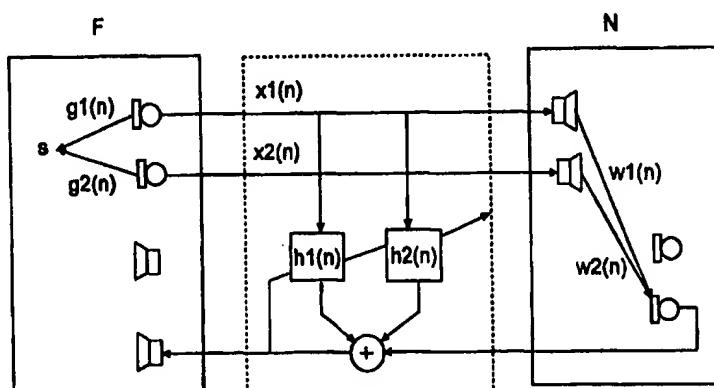


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(21) International Application Number: PCT/SE98/01858 (22) International Filing Date: 16 October 1998 (16.10.98) (30) Priority Data: 9703972-1        29 October 1997 (29.10.97)        SE (71) Applicant: TELIA AB (publ) [SE/SE]; Mårbackagatan 11, S-123 86 Farsta (SE). (72) Inventors: TILL, Ove; Sövsångarvägen 25, S-138 37 Ålta (SE). GÄNSLER, Tomas; Rudeboksvägen 91, S-226 55 Lund (SE). ENEROTH, Peter; Luzernvägen 1C, S-227 38 Lund (SE). (74) Agent: PRAGSTEN, Rolf; Telia Research AB, Vitsandsgatan 9, S-123 86 Farsta (SE).		(81) Designated States: EE, LT, LV, NO, European patent (AT, BE, CH, CY, DE, DK, ES, FI, FR, GB, GR, IE, IT, LU, MC, NL, PT, SE).  <b>Published</b> <i>Without international search report and to be republished upon receipt of that report.</i>

(54) Title: METHOD AND DEVICE AT STEREO ACOUSTIC ECHO CANCELLATION



## (57) Abstract

The present invention relates to a method and device at stereo acoustic echo cancellation. Acoustic echo cancellation in stereo is considerably more difficult than echo cancellation in mono, due to strong correlation between the stereo channels. This invention is based on utilization of a perceptual audio coder to reduce the correlation between the stereo channels, without introducing audible distortion. This will result in that the stereo canceller converges towards the correct echo paths and therefore gives a more stable echo cancellation which is not dependent on the transmission room (far-end). The core of the invention is that one can reduce the correlation in excess of that which is made by the audio coder, by modifying its decoder. Extra, uncorrelated (between the channels) noise is added (in the decoder) to such an extent that it is not audible, by information provided from the coder being used in combination with an estimated perceptual masking threshold. The solution consequently is flexible and does not require that the coding standard which is used is changed in any way. Only a small number of operations need to be included in the decoder.

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TITLE OF THE INVENTION: METHOD AND DEVICE AT STEREO  
ACCOUSTIC ECHO CANCELLATION

5 TECHNICAL FIELD

The present invention relates to echo cancellation in combination with signal coding.

TECHNICAL PROBLEM

10 Acoustic echo cancellation in stereo channels is a more difficult problem than corresponding mono case. This is due to the fact that each channel carries similar speech signals, which results in problems for the adaptive algorithm that is used. The fields of application for  
15 stereo cancellation is/is expected to be/ high-quality video conference systems and the field of tele-games. These fields, however, have different demands on quality, bandwidth etc.

20 In the mono case, NLMS (Normalized Least Mean Square Algorithm) is used exclusively, due to its robustness against noise and signal variations (none-stationariness). The disadvantage of this algorithm is that it has a convergence which is dependent on the spectral  
25 characteristics of incoming signals (far end). A strong (in time) auto correlated signal gives slow convergence and vice versa. In the stereo case, the speech signal is correlated in time, but also between respective channels which slows down the convergence speed for NLMS to such an  
30 extent that it will be useless. Echo cancellation then must be performed with some other kind of algorithm than NLMS. Essentially there are two types of algorithms to chose from, sub-band algorithms, or full length RLS (Recursive Least Square). These two of course have different  
35 advantages and disadvantages at implementation. The channel correlation also results in that there is no theoretical

estimate of echo paths which the echo canceller converges towards, but a lot of solutions which all are dependent on the transmitter room (far end), Figure 1. This results in an unstable echo cancellation, and the echo canceller  
 5 diverges with irregular intervals. To make the echo canceller converge in a stable way towards the correct echo paths, the stereo signals have to be modified before they reach the echo canceller as reference signals.

10 Stereo cancellation includes the following complex of problems:

- \* The echo paths  $w_1(n)$ ,  $w_2(n)$ , Figure 1, in the near end, N, which shall be estimated by AEC is not uniquely  
 15 indentifiable from measure data.
- \* The echo cancellation of the canceller is dependent on the variability of the channels,  $g_1(n)$ ,  $g_2(n)$  in the far end, F.

20 Assume that the signals from the microphone of the far end is given by, Figure 3,

$$x_i(n) = g_i(n) * s(n), \quad i=1,2$$

25 where  $s(n)$  is the source signal and  $g_i(n)$ ,  $i=1,2$  is the echo paths of the far end with the length M, "\*" describes convolution.

30 The residual echo/echoes after the echo canceller is

$$\begin{aligned} e(n) &= y(n) - h_{1,L}^T x_{1,L} - h_{2,L}^T x_{2,L} \\ y(n) &= h_{2,n}^T x_{1,N,(n)} + h_{2,n}^T x_{2,N,(n)} \\ h_{i,n} &= [h_{i,0} \dots h_{i,N-1}]^T \\ 35 \quad X_{i,N(n)} &= [X_{i(n)} \dots X_{i(n-N+1)}]^T \end{aligned}$$

$h_{iN}$ ,  $i=1,2$  is the real response of the length  $N$  from the near end, and  $h_{i,L}=1,2$  is the estimated response of the length  $L$ .

5 Minimization of the weighted least square criteria

$$J(n) = \sum_{l=1}^n \lambda^{n-l} |e(n)|^2, 0 < \lambda \leq 1$$

results in the solution of the linear equation system

10

$$\begin{bmatrix} R_{xx}(n) \hat{h}_{1,L} \\ \hat{h}_{2,L} \end{bmatrix} = r_{xx}(n)$$

where  $r_{xx}(n)$  is the estimated cross correlation vector, and

15  $R_{xx}(n)$  is the correlation matrix,

$$R_{xx}(n) = \sum_{l=1}^n \begin{bmatrix} x_{1,L}(l)x_{1,L}^T(l) & x_{2,L}(l)x_{1,L}^T(l) \\ x_{1,L}(l)x_{3,L}^T(l) & x_{3,L}(l)x_{2,L}^T(l) \end{bmatrix}$$

20 The problem at stereophonic echo cancellation is the conditional number for this matrix. Further has been shown

$L \geq M \Rightarrow R_{xx}(n)$  is singular  $n$

$L < M \Rightarrow R_{xx}(n)$  is poorly conditioned

25  $L \geq N \Rightarrow$  misalignment  $\varepsilon(n) \rightarrow 0, n \rightarrow \infty$

$L < N \Rightarrow$  misalignment  $\varepsilon(n) \rightarrow 0 \rightarrow \forall n$

where the misalignment is

30  $\varepsilon(n) = \|h - \hat{h}\|^2 / \|h\|^2$  and  $\hat{h} = \begin{bmatrix} \hat{h}_{1,L}^T & \hat{h}_{2,L}^T \end{bmatrix}^T$   $h = \begin{bmatrix} h_{1,L}^T & h_{2,L}^T \end{bmatrix}^T$

A poorly conditioned  $R_{xx}(n)$  increases the misalignment. Consequently there is a contradiction in the solution if

$L < M$  is better conditioned, on the other hand the misalignment is reduced if  $L \geq N$ , but practically is  $L < M = N$ . The solution of this misalignment is to reduce the correlation between the stereo channels.

5 The eigenvalues of the correlation matrix can be limited in downward direction by  $[1 - |\gamma(f)|^2]$ , where  $\gamma(f)$  is the coherence between the stereo channels. Misalignment therefore can be measured with the coherence function, which then serves as  
10 a measure of achieved decorrelation.

The present inventions therefore are intended to solve the above mentioned problems.

#### 15 PRIOR ART

Two important applications for stereo acoustic echo cancellation is high-quality video conference and tele games. In the future, also desk-top based conference systems will have a need for stereo acoustic echo  
20 cancellers (AEC). These systems have different demands on bandwidth, bit rate etc.

Stereo acoustic echo cancelling, however, has turned out to be more complicated than the mono channel case. This is due  
25 to that, in the two channel case, the signals are linearly depending, which results in convergence problems for the echo canceller. Because of the linear dependence between the channels, there are theoretically no unique solutions for the echo canceller to identify. Furthermore, all not  
30 unique solutions are depending on the echo paths at the far end of the connection, F, (far end). In real situations, however, the solution is not singular, but only poorly conditioned due to uncorrelated microphone noise and infinitely long impulse responses on the echo paths of the  
35 far end. The convergence degree of the NMLS-algorithm is to

a great extent depending on the number of the system conditions, so more sophisticated algorithms are needed at stereo acoustic echo cancellation.

5 Beside the utilization of more sophisticated algorithms, problems remain with unstable estimates of the echo paths. In order to stabilize the solution, the correlation between the stereo channels must be reduced without introduction of  
10 distorting distortion. Different solutions to solve this have been presented, but have been rejected for different reasons (see for instance M.M. Sondhi, D.R. Morgan and J.L. Gall: "Stereophonic acoustic cancellation - an overview of the fundamental Problem"; IEEE Signal Processing Letters, 2(8):148-151, 1955). The most promising solution at present  
15 is to distort the stereo channels non-linearly (for instance J. Benesty, R. Morgan and M.M. Sondhi: "A better understanding and an improved solution to the problem of stereophonic acoustic echo cancellation". IEE Trans. On Speech and Audio Processing. To appear; A short version can  
20 be found in Proc. of ICASSP 1997 pp 303-306) where half-wave rectified parts of the signal are added to the signal itself. This distortion does not destroy the stereophonic perception, but introduces noise which for the most part is inaudible, but can be registered depending on the extent of  
25 non-linearity.

At transmission of acoustic signals between parties in, for instance, a telecommunication, a certain part of the own sound is brought back and creates an echo. In most cases  
30 one wants to have this echo at least reduced to a level which is not disturbing. This is achieved by means of a so called echo canceller. The principle for these is that a part of the own signal is identified and subtracted from the received signal. It consequently is known to utilize  
35 echo cancellation in the mono case. At this previously known principles are utilized which i.a. are described in

the patent literature, for instance US 5668865, US 5664011, US 5610909. In the patent documents US 5661813, US 574545, US 5323459, US 5369554, US 5555310 and US 5513265, the problems of stereo acoustic echo cancellation are dealt with more specifically.

#### THE SOLUTION

The present invention relates to a method at stereo acoustic echo cancellation, where the echo is created on a connection for transmission of a stereo acoustic signal. The signal is coded on the transmitter side, F, and decoded on the receiver side, N. A perceptual audio coding is introduced. By perceptual coding is meant that the signal can consist of different frequencies which are transmitted at the same time, there one of these signals dominates over the other but gives no additional contribution to the received information. Furthermore, the side information of the coded signal is utilized. The echo after that can be identified and cancelled. By utilization of, for instance, MPEG-coding, perceptual coding which allows that the channel correlation between the stereo channels is reduced, is achieved. At frequencies over 2 kHz the perceptual coding is advantageous. Below 2 kHz, the side information can be utilized to further reduce the correlation. Each sub-band into which the signal is divided, indicates the utilization of the signal and the quantizer which is used at the coding. Quantizer is selected at the coding at which at an analysed segment of the signal is utilized. Further, a masking threshold is appointed which defines distortion levels which cannot be heard within the segment. The masking threshold is selected so that a just noticeable distortion is attained. Uncorrelated noise between the channels are added to the margin in the decoder, at which an improved echo cancellation can be achieved.



The invention further relates to a device at stereo acoustic echo cancellation. A sound registering equipment on the transmitter side, F, registers the signal which is coded in a coder, C, and transmitted to a decoder, D, on the receiver side, N. In the coder, C, a perceptual coding of the signal is performed. Side information in the coded signal is further utilized. For identification of the echo and the cancellation of this, a stereo acoustic echo cancellator, AEC, is utilized. Perceptual coding is performed in the coder, C, by utilization of, for instance, MPEG-coding for reduction of the channel correlation between the channels. The decoder analyses segment of the signal for deciding a masking threshold which defines inaudible distortion levels within the segment. The coder, C, further selects quantizer, dq. Appointment of masking threshold is made in the decoder in such a way that a margin to just noticeable distortion is attained. Uncorrelated noise, between the channels, is added by the decoder to the signal.

20

#### ADVANTAGES

The invention makes possible that methods for cancellation of echoes, on connections over which stereo transmissions are made, are executed. The introduction of the invention is possible without addition of extra equipment, which may be expensive. By utilization of perceptual coders/decoders a possibility is given to implement the solution on the decoder side, without the coder having need for knowing this. The solution further has the advantage that a good conditioning is attained, without introducing distortion, which can interfere with the communication.

30

#### DESCRIPTION OF FIGURES

Figure 1 illustrates microphone and loudspeaker near-end, N, respective far-end, F. Within the frame with the broken

35

line is the acoustic echo canceller (AEC), the stereo case. Only one of the back channels is shown.

Figure 2 illustrates the far-end room, stereo acoustic echo canceller, AEC, (stereo AEC) and perceptual audio coder, C/D (coder/decoder).

Figure 3 illustrates an MPEG-1 layer III decoder. The following designations have been used:

10	pi: PCM input
	af: Filter bank analysis
	md: MDCT
	sq: Scaling device and Quantizer
	hc: Huffman coding
15	mp: Multiplexer
	dm: Demultiplexer
	hd: Huffman decoding
	dd: Dequantizer and descaling device
	im: Inverse MDCT
20	sfb: Synthesis Filter Bank
	po: PCM output
	dt: decide masking thresholds
	si: side information
	di: Decoding of side information
25	b: MPEG layer III bit stream

Figure 4 illustrates the masking threshold. The dotted areas are masked by the tone. The sound pressure is indicated in dB. The frequency is indicated on a log scale. DN signifies decorrelating noise level. Q signifies quantizing noise level.

#### PREFERRED EMBODIMENT

In the following the invention is described on basis of the figures and the terms in them. Acoustic echo cancellation in stereo is considerably more difficult than echo

cancellation in mono, due to strong correlation between the stereo channels.

This invention is based on utilizing a perceptual audio  
5 coder to reduce the correlation between the stereo channels  
without introducing audible distortion. This will result in  
that the stereo canceller converges towards the correct  
echo paths and therefore gives a more stable echo  
cancellation which is not depending on the transmission  
10 room (far-end). The core of the invention is that one can  
reduce the correlation beyond that which the audio coder  
gives, by modifying its decoder. Extra uncorrelated  
(between the channels) noise is added (in the decoder) to  
such an extent that it is not audible, by the information  
15 from the coder being used in combination with an estimated  
perceptual masking threshold.

The solution consequently is flexible and does not require  
that the used coding standard is changed in any way. Only a  
20 small number of operations need to be included in the  
decoder.

The invention is based on that the distortion is introduced  
as noise addition to the speech signal without interfering  
25 with this. Further, the qualities of the speech/audio coder  
(for instance MPEG-coder) which is on the transmission  
channel, C/D, between the near-end and the far-end, is  
utilized. For the purpose, a perceptual audio coder, which  
introduces the effect that the channel correlation is  
30 reduced between the stereo channels, is utilized. The  
coherence will go down below 0,95 for frequencies over 2  
kHz with the MPEG Layer III coder. A coherence below 0,95  
is aimed at, to condition the solution which the echo  
canceller shall find. At frequencies below 2 kHz the  
35 coherence still is high, so further modification of the  
signal is necessary in the range below 2 kHz. For this

purpose, side information which is in the coded signal is utilized, without disturbing distortion being introduced. Within each sub-band of the signal which is decoded, the utilization of the signal is indicated, and which quantizer  
5 that the coder has utilized. The coder selects quantizer on basis of the amount of energy that is in the analysed segment of speech (or audio signal), and the so called masking threshold which indicates not audible distortion levels in the segment. Selection is made with knowledge of  
10 that there often is a margin to the just noticeable distortion level. The margin left is utilized by uncorrelated noise between the channels being added to the signal. By this measure, a coherence reduction is attained to find stable unique estimates of the echo paths in the  
15 near-end, N.

The most advanced part in the PMEG-1 standard is layer III, which typically compresses stereo sound up to 12 times without significant loss of quality of the sound. It is  
20 included in the standards such as H.310 audiovisual, broadband communications system, and H.323 visual telephone systems and equipment for local networks. Layer III coders usually also are utilized as high quality coders in World Wide Web (WWW).

25 The high compression is possible by removing parts in the signal which are not audible, or are lacking information for the ear. At simultaneous masking, larger frequency components will screen off the smaller ones in nearby  
30 frequency bands, whereas at temporary masking, i.e. components just before or after (in the time domain), a big sound component is screened off. The audio coder estimates the global masking threshold, the just noticeable distortion, as a function of frequency and time segment.

The sound decoder operates parallel with the global algorithm for estimation of the masking. The signal of the sound source is divided into 32 critically sampled down bandpass signals in a filter bank. In layer III the frequency selection is increased by each band pass signal being worked upon with a discrete cosine transform (MDCT). The length of the MDCT-window is signal dependent and is either 6 or 18, where the shorter window is utilized for transients in the sound source. The MDCT-components are scaled and quantized after the decompression. The key for noticing coder is that sufficient number of quantizing levels in each sub-band exist for keeping the introduced quantizing noise below the global masking threshold. The data redundancy is reduced by utilizing Huffman coding on the signal before it is transmitted in the channel.

When two signals are not identical, the introduced quantization noise in the two channels is almost independent. This will result in that the correlation between the channels is reduced. Decoding is essentially performed in the same way as the coding, but just the reverse.

The correlation between the channels are reduced even more if independent noise is added to the channels. Each of the DCT-bands cannot be optimally quantized due to big overhead. They are instead divided into five ranges with a defined number of quantizing levels. Define noise to mask relation (QMR) as the difference between the level of the quantizing noise and the level which is just audible in a given MDCT-band. After that, noise which is not audible can be added to the MDCT where QMR is positive. In the frequency ranges where the channel correlation need to be reduced to fulfil

35

$$\text{QMR}(j) > 0 \Rightarrow \tilde{X}_{\text{mdct}}^j = X_{\text{mdct}}^j + f(\text{QMR}(j)) \cdot v$$

$$\text{QMR}(j) < 0 \Rightarrow \tilde{X}_{\text{mdct}}^j = X_{\text{mdct}}^j$$

where  $X_{\text{mdct}}^j$  is the MDCT-component in band  $j$  and  $f(.)$  amplifies the noise component  $v$  which is added. A block  
5 implementing this channel decorrelation is added to the decoder just before the inverting of MDCT.

The global masking information is not accessible in the decoder, but thanks to the high frequency resolution of  
10 MDCT, a global masking estimate, the calculation complexity of which is simplified, is produced. Independent noise after that is added before the inverse MDCT in the MDCT-components which have sufficiently high SMR.

15 The invention is not restricted to the in the above described example of embodiment, or to the following patent claims, but may be subject to modifications within the frame of the idea of the invention.

## PATENT CLAIMS

1. Method at stereo acoustic echo cancellation, where the echo is created on a connection for transmission of a stereo acoustic signal, which signal is coded on the transmitter side (F), and decoded on the receiver side (N), characterized in that a perceptual audio coding is introduced, that side information in the coded signal is utilized, and that the echo can be identified and cancelled.
2. Method according to patent claim 1, characterized in that the perceptual coding is realized by, for instance, MPEG-coding, that the perceptual coding allows that the channel correlation is reduced between the stereo channels.
3. Method according to patent claim 1 and 2, characterized in that the perceptual coding with advantage is utilized at frequencies exceeding 2 kHz.
4. Method according to patent claim 1, characterized in that the side information preferably is utilized at frequencies up to 2 kHz.
5. Method according to patent claim 1 and 4, characterized in that, for respective sub-band in the signal, the utilization of the signal, and which quantizer that is used at the coding, is indicated.
6. Method according to patent claim 1, 4 and 5, characterized in that the quantizer is selected at the coding on basis of an analysed segment of the signal, and that a masking threshold, indicating

not audible distorsion levels within the segment, is selected.

7. Method according to patent claim 1, 4, 5 and 6,  
5 c h a r a c t e r i z e d in that the selection of the masking threshold is made so that a margin to a just noticeable distorsion is attained, and in that uncorrelated noise between the channels is added to the margin.  
10
8. Device at stereo acoustic echo cancellation, where a signal is registered by a sound registering equipment on the transmitter side, F, and the signal is coded in a coder (C) and transmitted on a connection to a decoder  
15 (D) on the receiver side (N), c h a r a c t e r i z e d in that the coder (C) is arranged to perform a perceptual coding of the signal, that side information in the coded signal is utilized, and that a stereo acoustic echo canceller is arranged to identify the echo  
20 and reduce it.
9. Device according to patent claim 8,  
c h a r a c t e r i z e d in that the perceptual coding is performed in the coder (C), and that, for instance,  
25 MPEG-coding is utilized for reduction of channel correlation between channels.
10. Device according to patent claim 8,  
c h a r a c t e r i z e d in that the stereo acoustic  
30 echo canceller (AEC) is arranged to analyse segment of the signal to appoint a masking threshold defining unaudible distorsion levels within the segment.
11. Device according to patent claim 8 or 10,  
35 c h a r a c t e r i z e d in that the coder (C) is arranged to select the quantizer.



12. Device according to patent claim 8, 9, 10, or 11,  
c h a r a c t e r i z e d in that the stereo acoustic  
echo canceller (AEC) is arranged to select the masking  
threshold so that a margin to a just noticeable  
5     distorsion is attained, that the stereo acoustic echo  
canceller (AEC) is arranged to add an uncorrelated noise  
between the channels, to the signal.

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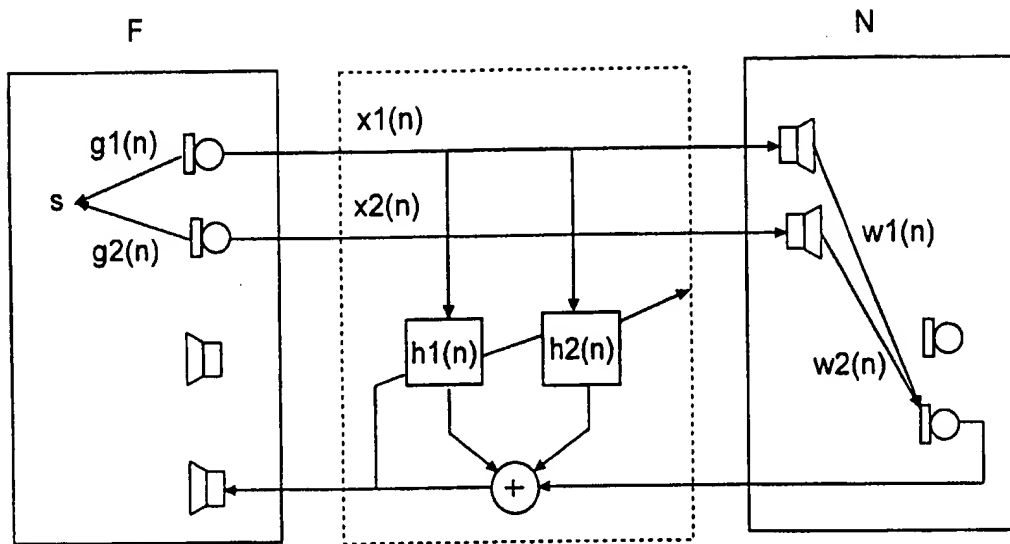


Figure 1

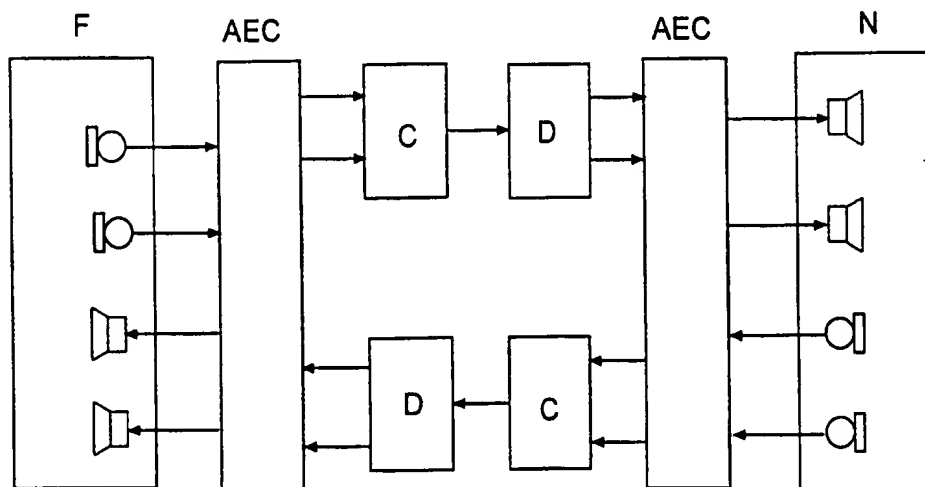


Figure 2

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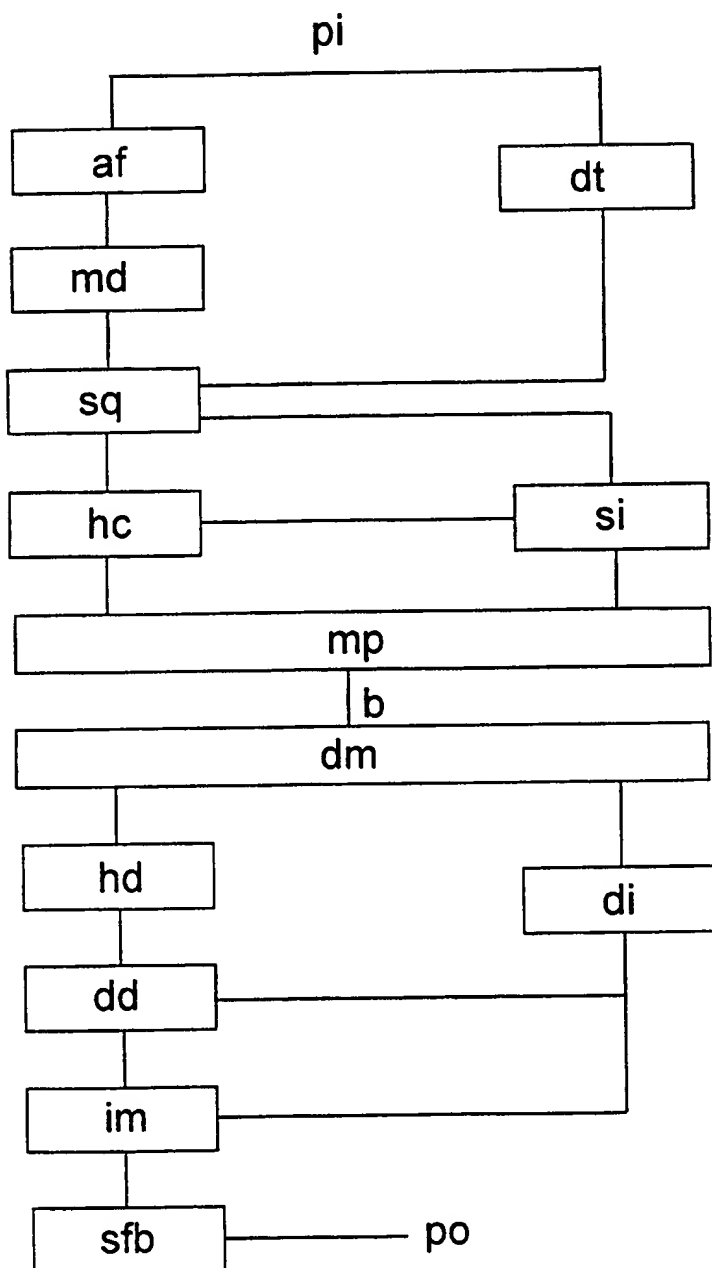


Figure 3

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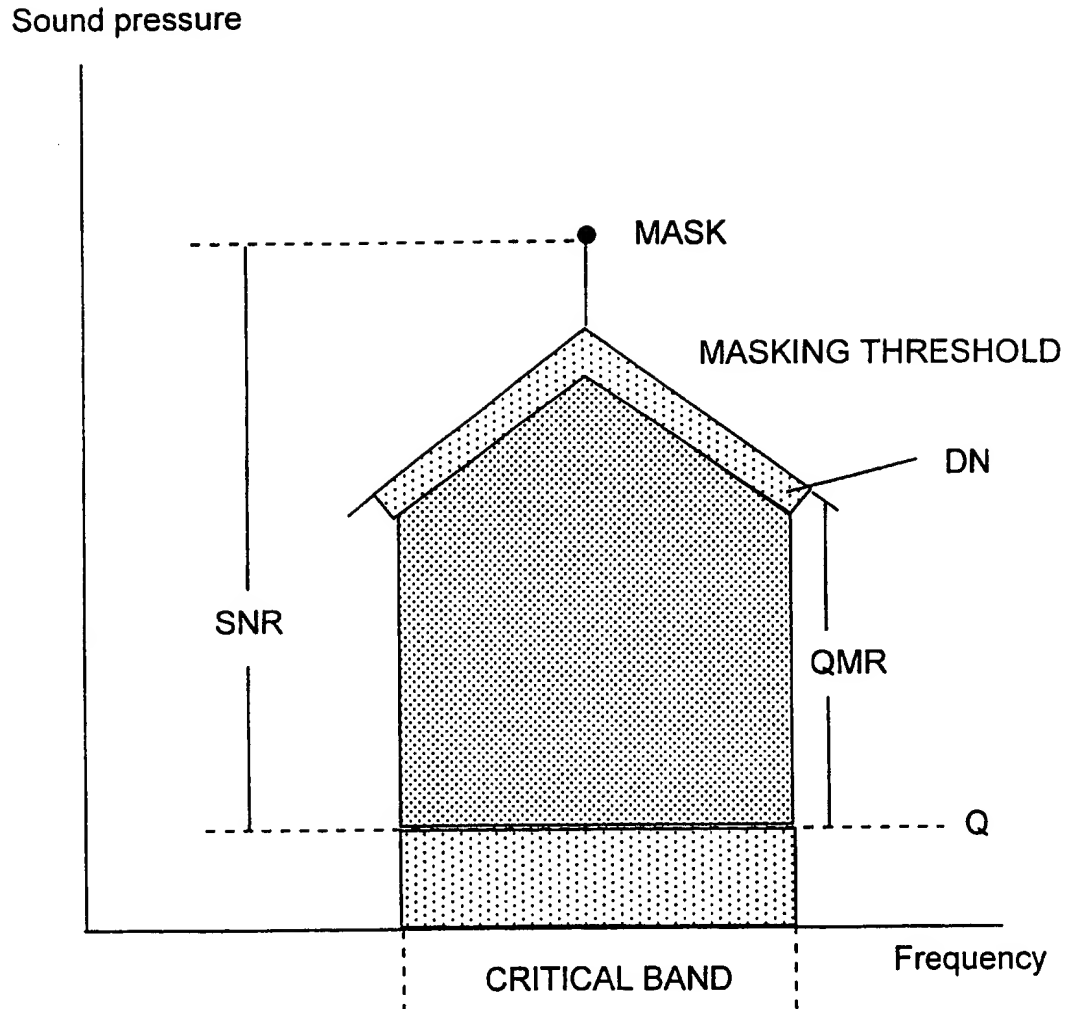


Figure 4

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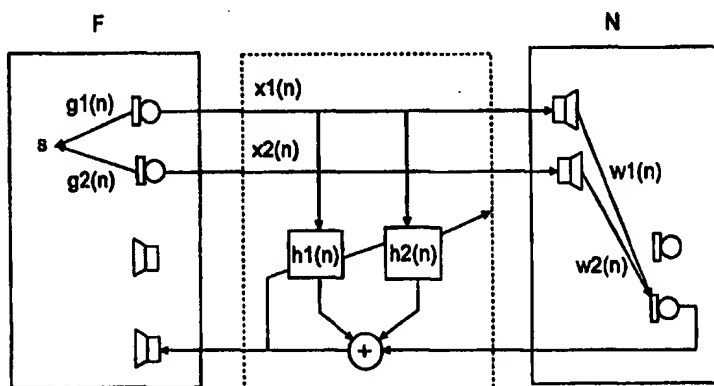
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(72) Inventors: TILL, Ove; Sövsångarvägen 25, S-138 37 Älta (SE). GÄNSLER, Tomas; Rudeboksvägen 91, S-226 55 Lund (SE). ENEROTH, Peter; Luzernvägen 1C, S-227 38 Lund (SE).			
(74) Agent: PRAGSTEN, Rolf; Telia Research AB, Vitsandsgatan 9, S-123 86 Farsta (SE).			

(54) Title: METHOD AND DEVICE AT STEREO ACOUSTIC ECHO CANCELLATION



(57) Abstract

The present invention relates to a method and device at stereo acoustic echo cancellation. Acoustic echo cancellation in stereo is considerably more difficult than echo cancellation in mono, due to strong correlation between the stereo channels. This invention is based on utilization of a perceptual audio coder to reduce the correlation between the stereo channels, without introducing audible distortion. This will result in that the stereo canceller converges towards the correct echo paths and therefore gives a more stable echo cancellation which is not dependent on the transmission room (far-end). The core of the invention is that one can reduce the correlation in excess of that which is made by the audio coder, by modifying its decoder. Extra, uncorrelated (between the channels) noise is added (in the decoder) to such an extent that it is not audible, by information provided from the coder being used in combination with an estimated perceptual masking threshold. The solution consequently is flexible and does not require that the coding standard which is used is changed in any way. Only a small number of operations need to be included in the decoder.

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BG	Bulgaria	HU	Hungary	MN	Mongolia	TT	Trinidad and Tobago
BJ	Benin	IE	Ireland	MR	Mauritania	UA	Ukraine
BR	Brazil	IL	Israel	MW	Malawi	UG	Uganda
BY	Belarus	IS	Iceland	MX	Mexico	US	United States of America
CA	Canada	IT	Italy	NE	Niger	UZ	Uzbekistan
CF	Central African Republic	JP	Japan	NL	Netherlands	VN	Viet Nam
CG	Congo	KE	Kenya	NO	Norway	YU	Yugoslavia
CH	Switzerland	KG	Kyrgyzstan	NZ	New Zealand	ZW	Zimbabwe
CI	Côte d'Ivoire	KP	Democratic People's Republic of Korea	PL	Poland		
CM	Cameroon	KR	Republic of Korea	PT	Portugal		
CN	China	KZ	Kazakhstan	RO	Romania		
CU	Cuba	LC	Saint Lucia	RU	Russian Federation		
CZ	Czech Republic	LI	Liechtenstein	SD	Sudan		
DE	Germany	LK	Sri Lanka	SE	Sweden		
DK	Denmark	LR	Liberia	SG	Singapore		
EE	Estonia						

## INTERNATIONAL SEARCH REPORT

International application No.

PCT/SE 98/01858

## A. CLASSIFICATION OF SUBJECT MATTER

IPC6: H04B 3/20

According to International Patent Classification (IPC) or to both national classification and IPC

## B. FIELDS SEARCHED

Minimum documentation searched (classification system followed by classification symbols)

IPC6: H04B

Documentation searched other than minimum documentation to the extent that such documents are included in the fields searched

SE,DK,FI,NO classes as above

Electronic data base consulted during the international search (name of data base and, where practicable, search terms used)

WPI, JAPIO

## C. DOCUMENTS CONSIDERED TO BE RELEVANT

Category*	Citation of document, with indication, where appropriate, of the relevant passages	Relevant to claim No.
X	US 5661813 A (SUEHIRO SHIMAUCHI ET AL), 26 August 1997 (26.08.97), column 15, line 7 - column 19, line 24 --	1-12
A	US 5323459 A (AKIHIRO HIRANO), 21 June 1994 (21.06.94), see the whole document --	1-12
A	US 5513265 A (AKIHIRO HIRANO), 30 April 1996 (30.04.96), see the whole document --	1-12
A	US 5555310 A (SHIGENOBU MINAMI ET AL), 10 Sept 1996 (10.09.96), see the whole document -- -----	1-12

☐ Further documents are listed in the continuation of Box C.☒ See patent family annex.

\* Special categories of cited documents:

"A" document defining the general state of the art which is not considered to be of particular relevance

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"&amp;" document member of the same patent family

Date of the actual completion of the international search

29 April 1999

Date of mailing of the international search report

05-05-1999

Name and mailing address of the ISA/

Swedish Patent Office

Box 5055, S-102 42 STOCKHOLM

Facsimile No. +46 8 666 02 86

Authorized officer

Bengt Romedahl

Telephone No. +46 8 782 25 00

**INTERNATIONAL SEARCH REPORT**  
Information on patent family members

07/04/99

International application No.  
**PCT/SE 98/01858**

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